

mation and converts the text response to an audio signal in accordance with the call origination language.

BRIEF DESCRIPTION OF THE DRAWINGS

[0014] The subject matter which is regarded as the invention is particularly pointed out and distinctly claimed in the claims at the conclusion of the specification. The foregoing and other objects, features, and advantages of the invention are apparent from the following detailed description taken in conjunction with the accompanying drawings in which:

[0015] FIG. 1 illustrates one example of a block diagram in accordance with the present invention for synthesizing responses using a MLTTS synthesizer in a call center system; and

[0016] FIG. 2 is a flowchart illustrating one method of the present invention shown in FIG. 1.

[0017] The detailed description explains the preferred embodiments of the invention together with advantages and features, by way of example with reference to the drawings.

DETAILED DESCRIPTION OF THE INVENTION

[0018] Turning now to the drawings in greater detail, it will be seen that in FIG. 1 there is a block diagram depicting aspects of a runtime system to interactively synthesize responses to a caller using a multi-language text-to-speech (MLTTS) synthesizer. In the exemplary embodiments, the MLTTS is used in a call center environment and provides outbound audio to a caller in at least one of the same language and dialect as that of the caller.

[0019] FIG. 1 shows an exemplary call center configuration. The configuration of FIG. 1 is illustrative rather than limiting of the teachings herein.

[0020] As shown in FIG. 1, a caller 100, using either a wireless phone or a wired phone, places a call to a call center whose purpose is usually to distribute the telephone calls to available customer service representatives, referred to herein as “call handlers.” The call center will distribute the incoming calls using any one of numerous well known automatic call distribution techniques to one node in the call center wherein the call is handled by a call handler in the call handler node 210. In this particular embodiment, the invention shows the distribution of the calls occurring via a public switch telephone network (PSTN) 110. This invention is not limited in this way, however, and applies as well as when other kinds of networks are employed, including voice-over-IP networks, cellular telephone networks, satellite networks, emergency networks, private corporation networks, and the like.

[0021] The PSTN 110 sends the input of the call into an Interactive Voice Response (IVR) platform 120. The IVR platform 120 includes a database 121 and is capable of accepting a combination of voice telephone input and touch-tone keypad selection but is not limited to this combination. In one embodiment, the database 121 will include both area and world telephone codes of telephone numbers and the corresponding language associated with the area and world telephone codes. Information, including the caller's audio message, from the IVR platform 120 is sent to a media splitter 130. The media splitter 130 is also capable of sending information back to the IVR platform 120 and then in turn to the caller 100 through the PSTN 110. The media splitter 130 receives inbound calls from the PSTN 110 and

sets up a connection with the inbound audio channel that has a telephone adapter 220 connected to a speaker 230 or headset so that the call handler can listen to the caller 100. The media splitter 130 also routes the information to the call handler and simultaneously opens a Voice Extensible Markup Language (XML) browser 140 session. The Voice XML 140 receives its information from a workstation and graphical user interface (GUI) 240. When the call handler receives a call, the call handler listens to the caller's 100 audio signal and replies to the caller 100 by typing the response to the caller 100 into a workstation with a graphical user interface (GUI) 240. The output from the GUI 240 is used as input into the Voice XML browser 140.

[0022] The Voice XML browser 140 receives information from the workstation with the GUI 240, whereby the call handler, after listening to the incoming audio on a speaker 230, responds to the caller 100 by entering a response message through the GUI at the workstation 240. The Voice XML browser 140 sends and receives signals and information to a voice server 150. The voice server 150 upon receiving the response message sends the response message to a text to speech (MLTTS) synthesizer 160. The text to speech synthesizer 160 processes the response message in accordance with information received from the IVR platform 120 and database 121 and sends audio signals back to the caller 100 by routing the information through the media splitter 130 to the IVR platform 120 through the telephone network 110 to the caller 100. In other words, the MLTTS synthesizer 160 synthesizes the outgoing audio so that the output is in the native language and accent of the caller 100 so that the outgoing voice sounds familiar to the caller 100. The preferred method uses a very high quality synthesizer 160, such as IBM Web Sphere Voice™ server, to synthesize responses to the caller's queries.

[0023] In an alternate embodiment, the database 121 sends the desired language response information directly to the MLTTS synthesizer 160. With the above setup in place, a call handler 250 is able to interactively respond to a caller 100 via a speech synthesizer 160. The IVR platform 120 is capable of providing the speech synthesizer 160 the information to select the correct language based on the incoming phone number and a corresponding database 121. After initializing the appropriate MLTTS synthesizer 160 based on the incoming call (for example, a synthesizer for one of the United States, the United Kingdom, or other language) responses are provided to the caller 100 in the caller's language.

[0024] One example of the incoming phone number being mapped to a language could be as follows: 1 800 XXX XXX2—can be mapped to United States English whereas 1 800 XXX XXX3—can be mapped to United Kingdom English.

[0025] Referring to FIG. 2, there is shown a flow diagram of one embodiment in accordance with FIG. 1. One scenario is as follows. A caller places a call 300. The network receives the call and distributes 310 the call to the IVR platform. The platform then determines and assigns a language based on the incoming caller's telephone number after looking up and matching the information in a database 320. The IVR 120 sends the information and signal to the media splitter 130 so that the splitter can simultaneously initialize a Voice XML Browser 370 and rings a free call handler's extension 350 and assigns 340 the inbound audio to that extension. At this point, the call handler will see 360 a screen pop-up at a